## IN THE CLAIMS

Please amend the claims as follows:

- 1. (Currently Amended) A device for use with a voice gateway coupled in a network adapted to transmit network packets that meet a minimum packet switched network protocol, comprising:
- a WB Wide Band (WB) telephone adapted to convert sound into sound signals that capture a wideband bandwidth of the sound that includes a range of 200 Hz to 5 kHz;

an encoder coupled to receive the sound signals and to encode them as voice data bits;

- a packetizer for packetizing groups of the voice data bits into intermediate packets which do not meet the minimum packet switched network protocol; and
- a modem adapted to establish a first circuit switched connection with the voice gateway, and coupled to transmit the intermediate packets through the first connection.
- 2. (Original) The device of claim 1, further comprising: a decoder coupled to receive other voice data bits through the connection and the modem.
- 3. (Currently Amended) The device of claim 1, wherein the modem is a DSVD Digital Simultaneous Voice and Data (DSVD) modem.
- 4. (Currently Amended) The device of claim 1, wherein

  the first connection has a rated capacity of 28.8 kbps the wideband bandwidth of the

  sound is within a range of 200 Hertz (Hz) to 5 Hz.
- 5. (Currently Amended) The device of claim 1, wherein

  the bandwidth includes a range of 150 Hz to 7.1 kHz the sound signals are encoded to
  capture the wideband signal range and the first circuit switched connection only supports a
  bandwidth lower than the wideband bandwidth.
- (Original) The device of claim 1, wherein the encoder encodes at a rate of at least 16 kbps.

## 7. (Currently Amended) A device comprising:

means for establishing a first circuit switched telephone connection with a first device at a first endpoint of a network capable of transmitting network data packets which meet a minimum packet switched network protocol;

means for converting sound into sound signals;

means for encoding the sound signals into voice data bits at a rate of at least 16 kbps; means for packetizing groups of the voice data bits into intermediate packets which do not meet the minimum packet switched network protocol; and

means for transmitting the intermediate packets through the first circuit switched telephone connection.

- 8. (Original) The device of claim 7, further comprising:
  means for multiplexing additional data with the voice data bits prior to transmitting.
- 9. (Original) The device of claim 7, further comprising:

  means for receiving through the first connection return intermediate packets;

  means for depacketizing the return intermediate packets to derive return voice data

  bits;

  means for decoding the return voice data bits to produce return sound signals; and

  means for producing a return sound from the return sound signals.
- 10. (Currently Amended) The device of claim 7, wherein

  the first connection has a rated capacity of 28.8 kbps the sound signals are encoded into voice signals at a wideband encoding rate and transmitted over the first circuit switched telephone connection at a rate that is less than the wideband encoding rate.
- 11. (Currently Amended) A gateway comprising:
  - a network interface for coupling to a network; and
- a processor coupled with the network interface, wherein the processor is adapted to establish a first circuit switched telephone connection with a modem;

establish a second packet switched network connection through a network with a device at an endpoint of the network;

receive through the first connection a stream of intermediate <u>packet switched network</u> packets that <u>and do not include headers necessary for transport over the packet switched</u>

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network and include voice data bits which represent sound that has been encoded at a rate of at least 16 kbps;

add packet headers to the intermediate packets to form network packets; and transmit the network packets through the second connection.

- 12. (Original) The gateway of claim 11, wherein the processor is further adapted to receive first dialing information,
  wherein the first dialing information is used to establish the second connection.
- 13. (Original) The gateway of claim 11, wherein the packet headers include at least one of IP type headers, UDP type headers and RTP type headers.
- 14. (Currently Amended) The gateway of claim 11, wherein

  the first connection supports a data transmission at a rate not exceeding 28.8 kbps the

  voice data bits have been encoded at a high bandwidth rate and the first circuit switched

  telephone connection only supports a lower bandwidth rate.
- 15. (Original) The gateway of claim 11, wherein
  the stream includes non-voice data bits,
  and the processor is further adapted to:
  demultiplex the voice data bits from the non-voice data bits in the stream prior to
  adding the headers.
- 16. (Currently Amended) An adapter for an analog wideband telephone to communicate with a voice gateway coupled in a network adapted to transmit network packets that meet a minimum packet switched network protocol, the adapter comprising:

an analog to digital converter for digitizing voice signals received from the analog wideband telephone;

an encoder coupled to receive the digitized voice signals and to encode them as voice data bits;

a packetizer for packetizing groups of the voice data bits into intermediate packets which do not meet the minimum packet switched network protocol; and

a modem adapted to establish a first circuit switched connection with the voice gateway, and adapted to be coupled to transmit the intermediate packets through the first connection.

- 17. (Original) The adapter of claim 16, further comprising:
- a depacketizer for depacketizing return intermediate packets which do not meet the minimum protocol to produce return voice data bits;
- a decoder for decoding the return voice data bits to produce a digital return voice signal; and
- a digital to analog converter for converting the digital return voice signal into an analog signal for the analog wideband telephone.
- 18. (Currently Amended) A gateway comprising:
  - a network interface for coupling to a network; and
- a processor coupled with the network interface, wherein the processor is adapted to establish a first packet switched network connection through a network with a device at an endpoint of the network;

establish a second circuit switched telephone connection with a modem;

receive through the first connection a stream of network packets that transport voice data bits that represent sound which has been encoded at a rate of at least 16 kbps;

strip packet headers from the network packets to yield intermediate packets that do not conform with a minimum packet switched network protocol; and transmit the yielded intermediate packets through the second connection.

- 19. (Original) The gateway of claim 18, wherein the processor is further adapted to: receive second dialing information, wherein the second dialing information is used to establish the second connection.
- 20. (Original) The gateway of claim 18, wherein the packet headers include at least one of IP type headers, UDP type headers and RTP type headers.
- 21. (Original) The gateway of claim 18, wherein the first connection supports a data transmission at a rate not exceeding 28.8 kbps.

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## 22. (Original) A device comprising:

means for establishing a first packet switched network connection through a network with a device at an endpoint of the network;

means for establishing a second circuit switched telephone connection with a modem; means for receiving through the first connection a stream of network packets that transport voice data bits that represent sound which has been encoded at a rate of at least 16 kbps;

means for stripping packet headers from the network packets to yield intermediate packets; and

means for transmitting the yielded intermediate packets through the second connection.

23. (Original) The device of claim 22, further comprising:

means for receiving second dialing information,

wherein the second dialing information is used by the means for establishing the second connection.

- 24. (Original) The device of claim 22, wherein
- the packet headers include at least one of IP type headers, UDP type headers and RTP type headers.
- 25. (Original) The device of claim 22, wherein the first connection supports a data transmission at a rate not exceeding 28.8 kbps.
- 26. (Original) A device comprising:

means for establishing a circuit switched connection with a voice gateway coupled in a network which is adapted to transmit network packets that meet a minimum protocol;

means for receiving voice signals from an analog wideband telephone, the voice signals encoding sound in a range of at least 200 Hz to 5 kHz;

means for digitizing the received voice signals;

means for encoding the digitized voice signals as voice data bits at a rate of at least 16 kbps;

means for packetizing groups of the voice data bits into intermediate packets which do not meet the minimum protocol; and

means for transmitting the intermediate packets through the circuit switched connection.

27. (Original) The device of claim 26, further comprising:

means for depacketizing return intermediate packets which do not meet the minimum protocol to produce return voice data bits;

means for decoding the return voice data bits to produce a digital return voice signal; and

means for converting the digital return voice signal into an analog signal.

28. (Currently Amended) An article comprising: a storage medium, said storage medium having stored thereon instructions, that, when executed by at least one device, result in:

establishing a first circuit switched telephone connection with a modem;

establishing a second packet switched network connection through a network with a device at an endpoint of the network;

receiving through the first connection a stream of intermediate packets that include voice data bits which represent sound that has been encoded at a rate of at least 16 kbps and do not conform to a minimum packet switched network protocol;

adding packet headers to the intermediate packets to form network packets that conform with the packet switched network protocol; and

transmitting the network packets through the second connection.

- 29. (Original) The article of claim 28, wherein executing the instructions further results in: receiving first dialing information, wherein the first dialing information is used to establish the second connection.
- 30. (Original) The article of claim 29, wherein the packet headers include at least one of IP type headers, UDP type headers and RTP type headers.
- 31. (Original) The article of claim 30, wherein the first connection supports a data transmission at a rate not exceeding 28.8 kbps.

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30. 32. (Currently Amended) The article of claim 32 28, wherein

the stream includes non-voice data bits,

and the instructions further result in:

demultiplexing the voice data bits from the non-voice data bits in the stream prior to adding the headers.

31. 33. (Currently Amended) An article comprising: a storage medium, said storage medium having stored thereon instructions, that, when executed by at least one device, result in:

establishing a first packet switched network connection through a network with a device at an endpoint of the network;

establishing a second circuit switched telephone connection with a modem;

receiving through the first connection a stream of network packets that transport voice data bits that represent sound which has been encoded at a rate of at least 16 kbps and do not conform to a minimum packet switched network protocol;

stripping packet headers from the network packets to yield intermediate packets that conform with the packet switched network protocol; and

transmitting the yielded intermediate packets through the second connection.

32. 34. (Currently Amended) The article of claim 31 33, wherein executing the instructions further results in:

receiving second dialing information,

wherein the second dialing information is used to establish the second connection.

- 33. 35. (Currently Amended) The article of claim 31 33, wherein the packet headers include at least one of PPP type headers, IP type headers, UDP type headers and RTP type headers.
- 34. 36. (Currently Amended) A method comprising:

establishing a first circuit switched telephone connection with a first device at a first endpoint of a network capable of transmitting data packets which meet a minimum protocol; converting sound into sound signals;

encoding the sound signals into voice data bits at a rate of at least 16 kbps;

packetizing groups of the voice data bits into intermediate packets which do not meet the minimum protocol; and

transmitting the intermediate packets through the first connection.

- 35. 37. (Currently Amended) The method of claim 34 36, wherein the first connection supports a data transmission at a rate not exceeding 28.8 kbps.
- 36. 38. (Currently Amended) The method of claim 34 36, further comprising: multiplexing additional data with the voice data bits prior to transmitting.
- 37. 39. (Currently Amended) The method of claim 34 36, wherein the bandwidth includes a range of 150 Hz to 7.1 kHz.
- 38. 40. (Currently Amended) The method of claim 34 36, further comprising: receiving through the first connection return intermediate packets; depacketizing the return intermediate packets to derive return voice data bits; decoding the return voice data bits to produce return sound signals; and inputting the return sound signals into a speaker to produce a return sound in a second bandwidth that includes a range of 1 kHz to 5 kHz.
- 39. 41. (Currently Amended) A method comprising:
  establishing a first circuit switched telephone connection with a modem;
  establishing a second packet switched network connection through a network with a
  device at an endpoint of the network;

receiving through the first connection a stream of intermediate packets that include voice data bits which represent sound that has been encoded at a rate of at least 16 kbps and do not conform with the packet formatting required for sending packets over the second packet switched network;

adding packet headers to the intermediate packets to form network packets that conform with the packet formatting required for sending packets over the second packet switched network; and

transmitting the network packets through the second connection.

40. 42. (Currently Amended) The method of claim 39 41, further comprising:

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receiving first dialing information,
wherein the first dialing information is used to establish the second connection.

- 41. 43. (Currently Amended) The method of claim 39 41, wherein the packet headers include at least one of IP type headers, UDP type headers and RTP type headers.
- 42. 44. (Currently Amended) The method of claim 39 41, wherein the first connection supports a data transmission at a rate not exceeding 28.8 kbps.
- 43. 45. (Currently Amended) The method of claim 39 41, wherein the stream includes non-voice data bits, and further comprising:

  demultiplexing the voice data bits from the non-voice data bits in the stream prior to adding the headers.
- [44-] 46. (Currently Amended) A method comprising:
  establishing a first packet switched network connection through a network with a
  device at an endpoint of the network;

establishing a second circuit switched telephone connection with a modem that does not support transmission at a widehand sound rate;

receiving through the first connection a stream of packets that transport voice data bits that represent sound encoded at a wideband sound rate which has been encoded at a rate of at least 16 kbps includes packet headers required for transporting the packets across the packet switched network;

stripping the packet headers from the network packets to yield intermediate packets that no longer include the headers necessary for transporting the packets over the packet switched network; and

transmitting the intermediate packets through the second connection.

45. 47. (Currently Amended) The method of claim [44] 46, further comprising: receiving second dialing information regarding the modem, wherein the second dialing information is used to establish the second connection.

- 46. 48. (Currently Amended) The method of claim [44] 46, wherein the packet headers include at least one of IP type headers, UDP type headers and RTP type headers.
- 47. 49. (Currently Amended) The method of claim [44] 46, wherein the second connection supports a data transmission at a rate not exceeding 28.8 kbps.
- 48. 50. (Currently Amended) A method comprising:
  establishing a circuit switched connection with a voice gateway coupled in a network
  which is adapted to transmit network packets that meet a minimum protocol;

  receiving voice signals from an applied wideband telephone, the voice signals

receiving voice signals from an analog wideband telephone, the voice signals encoding sound in a range of at least 200 Hz to 5 kHz;

digitizing the received voice signals;
encoding the digitized voice signals as voice data bits at a rate of at least 16 kbps;
packetizing groups of the voice data bits into intermediate packets which do not meet
the minimum protocol; and

transmitting the intermediate packets through the circuit switched connection.

49. 51. (Currently Amended) The method of claim 48 50, further comprising:
depacketizing return intermediate packets which do not meet the minimum protocol to
produce return voice data bits;

decoding the return voice data bits to produce a digital return voice signal; and converting the digital return voice signal into an analog signal.